

## **“Sound Editing and Compositing from Variable Acoustic Sources”**

UROP Narrative Report by William Brueggemann

April 25, 2017

In the spring semester of 2017, I collaborated with my faculty mentor Dr. Justin Rubin to assemble a 65-70 minute album of his original music. Recorded in different locations, acoustic spaces and to varying degrees of fidelity, I was tasked with not only editing these recordings but through the utilization of modern digital recording software, altering each track to best achieve a unified aesthetic and a cohesive sound. This process proved to be both an incredible challenge as well as a profound learning experience for me as a budding audio engineer.

When I first set out to prepare for the research component of this grant I suspected that I would, through careful study and exhaustive exploration of existing texts, gain the majority of applicable skills required for a project like this. What I quickly learned was that despite my prior experience in this field and the many excellent resources at my disposal, there was still so much left for me to learn. This project pushed me to the limits of what I had thought possible and required me to take chances and experiment outside of what would be conventional in the realm of audio recording.

In November of 2016 I assisted Dr. Rubin in the engineering of a live recording on pianist Solon Pierce in Weber Music Hall. Though I had been a part of recordings before, this was the first time I had the opportunity to record using professional equipment and to record a musician of that caliber. There were several issues that came up during this recording session that involved a certain kind of ambient sound from the hall itself as the wood in the ceiling began

to creak in response to heat from the sunlight. We attempted to adjust microphone placement and wait for moments of quiet to begin recording.

This led to a series of recordings that captured excellent playing but an array of audio issues from pops and clips from the room as well as human noise. While the fidelity of these tracks was quite good, I knew that it would require major editing and audio production “wizardry” to overcome these substantial issues as well as maintaining the best performances from each take. From this session, I took the unbroken WAV file which was over 2 hours in length and edited together bits and pieces from each take to assemble 3 finished pieces each only 5-8 minutes in length. These edits were a challenge because of all the popping sounds from Weber Hall as well as the fact that there were 4 mic placements and numerous takes of each piece.

When the editing was complete, I worked with Dr. Rubin to address any notes he had in terms of performance and fidelity issues. I then went on to edit the many other recordings which he had previously accrued from players and colleagues around the U.S. and beyond. The challenge with these edits were the result of a lack of consistency between acoustic spaces, microphone placements, quality of engineering and number of performances (“takes”) of each piece. This meant that each session required different solutions to ensure that the best sound possible would be the result. In some cases, I employed digital equalization to boost or reduce certain frequencies that contributed to white noise or distortion. In other cases, I added some reverberation synthesis or “Reverb” to the pieces that were recorded in a smaller space in order to match them with the sound of Weber hall.

Much of this work was incredibly tedious and required hours of trial and error to uncover each solution. What most surprised me was that the information I had been learning to prepare

for this (which included many conversations with UMD's own audio expert Donald Schraufnagel) was for the most part, unrelated to the specific issues I encountered in mixing and mastering these recordings. The largest and most insightful component of "research" came from my own hours of experimentation and invaluable experience of working on the actual music files. Much of what I had learned, such as the function of compression, limiting, equalization, microphone usage and the study of acoustics, was still very important to my development as a thoughtful musician and producer but I was surprised at how much simply required hands-on experience.

After all of the music was assembled and mixed, I performed the final step of mastering the finished audio tracks. This step involved the "normalization" process which simply means ensuring that each piece has an appropriate amount of dynamic range while still maintaining relative consistency between tracks and within a single piece. I experimented with many different processes for mastering including the use of limiters and compressors which aim to equalize the differences in volumes within a single sound-wave recording. I used the VST plug-in "Elephant" by Voxengo to apply some subtle limiting to certain tracks to boost their volume, optimizing them for a multitude of platforms. In some cases, I opted to adjust volumes manually using "volume envelopes" to ensure that there was no loss in fidelity or timbre.

In the end, I have assembled 18 finished tracks of music all from different sources and refined them to have a complimentary (if not identical) acoustical and musical aesthetic. I am proud of the results of this research project, especially considering how far these recordings have come in terms of quality. The experiences and lessons I have learned are invaluable to me and will hopefully build the essential foundations of my career as a musician, composer and audio production engineer.



1.1.00  
0:00.000

513.000  
17:04.000

1025.1.00  
34:08.000

1537.1.00  
51:12.000

2049.1.00  
1:08:16.000

2561.1.00  
1:25:20.000

3073.1.00  
1:42:24.000

3585.1.00  
1:59:28.000

4097.1.00  
2:16:32.000

4609.1.00  
2:33:36.000

5121.1.00  
2:50:40.000

CLOSE MIC

0.00dB center

-60

-54

-48

-42

-36

-30

-24

-18

-12

-6

-inf

2 Close Mic.L

0.00dB 100%L

-60

-54

-48

-42

-36

-30

-24

-18

-12

-6

-inf

3 Close Mic.R

0.00dB 100%R

-60

-54

-48

-42

-36

-30

-24

-18

-12

-6

-inf

ROOM MIC

0.00dB center

-60

-54

-48

-42

-36

-30

-24

-18

-12

-6

-inf

5 Room Mic.L

0.00dB 100%L

-60

-54

-48

-42

-36

-30

-24

-18

-12

-6

-inf

6 Room Mic.R

0.00dB 100%R

-60

-54

-48

-42

-36

-30

-24

-18

-12

-6

-inf

Audio 1\_06.L.aif

Audio 1\_06.R.aif

Audio 2\_06.L.aif

Audio 2\_06.R.aif

CLOSE MIC [input] Left

638.224 / 21:14.620 [Stopped]

Rate: 1.0 BPM 120 4/4 Selection: 1.1.00 1.1.00 0.0.00

MASTER

-inf

-inf

12

6

0

6

12

18

24

30

36

42

-inf

-inf

CLOSE MIC

-inf

-6

-18

-30

-42

-54

-inf

2 Close Mic.L

-inf

-6

-18

-30

-42

-54

-inf

Close Mic.R

-inf

-6

-18

-30

-42

-54

-inf

ROOM MIC

-inf

-6

-18

-30

-42

-54

-inf

5 Room Mic.L

-inf

-6

-18

-30

-42

-54

-inf

Room Mic.R

-inf

-6

-18

-30

-42

-54

-inf

Mixer

Routing Matrix

Windows taskbar with icons for File Explorer, Firefox, Chrome, Word, Music, and other applications. System clock shows 10:19 PM 5/3/2017.



CLOSE MIC

0.00dB center

2 CML

0.00dB 100%L

3 CMR

0.00dB 100%R

ROOM MIC

-2.27dB center

5 RML

0.00dB 75%L

6 RMR

0.00dB 75%R

ROOM MIC [input] Left

1.4.34 / 0:01.669 [Stopped]

Rate: 1.0 BPM 120 4/4 Selection: 1.4.55 208.3.27 204.2.72

MASTER

0.00dB center

CLOSE MIC

0.00dB center

2 CML

0.00dB 100%L

CMR

0.00dB 100%R

ROOM MIC

-2.27dB center

5 RML

0.00dB 75%L

RMR

0.00dB 75%R

7

0.00dB center

1.00 17.1.00 33.1.00 49.1.00 65.1.00 81.1.00 97.1.00 113.1.00 129.1.00 145.1.00 161.1.00 177.1.00 193.1.00 209.1.00

0:00.000 0:32.000 1:04.000 1:36.000 2:08.000 2:40.000 3:12.000 3:44.000 4:16.000 4:48.000 5:20.000 5:52.000 6:24.000 6:56.000

Audio 1 ... Audio 1\_06.L.aif Audio 1\_06.L.aif Audio 1\_06.L.aif Audio 1\_06.L.aif

Audio 1 ... Audio 1\_06.R.aif Audio 1\_06.R.aif Audio 1\_06.R.aif Audio 1\_06.R.aif

Audio 2 ... Audio 2\_06.L.aif Audio 2\_06.L.aif Audio 2\_06.L.aif Audio 2\_06.L.aif

Audio 2 ... Audio 2\_06.R.aif Audio 2\_06.R.aif Audio 2\_06.R.aif Audio 2\_06.R.aif

1.4.34 / 0:01.669 [Stopped]

Rate: 1.0 BPM 120 4/4 Selection: 1.4.55 208.3.27 204.2.72

MASTER

0.00dB center

CLOSE MIC

0.00dB center

2 CML

0.00dB 100%L

CMR

0.00dB 100%R

ROOM MIC

-2.27dB center

5 RML

0.00dB 75%L

RMR

0.00dB 75%R

7

0.00dB center

Windows taskbar with icons for File Explorer, Firefox, Chrome, Word, Music, and other applications. System clock shows 10:17 PM 5/3/2017.



1.1.00 257.1.00 513.1.00 769.1.00 1025.1.00 1281.1.00 1537.1.00 1793.1.00 2049.1.00 2305.1.00 2561.1.00 2817.1.00 3073.1.00 3329.1.00 3585.1.00

0:00.000 8:32.000 17:04.000 25:36.000 34:08.000 42:40.000 51:12.000 59:44.000 1:08:16.000 1:16:48.000 1:25:20.000 1:33:52.000 1:42:24.000 1:50:56.000 1:59:28.000

CLOSE MIC

2. CML

3. CMR

ROOM MIC

5. RML

6. RMR

CML [input] Left

745.3.95 / 24:49.473 [Stopped]

Rate: 1.0 BPM 120 4/4 Selection: 1.4.55 208.3.27 204.2.72

MASTER

CLOSE MIC

CML

CMR

ROOM MIC

RML

RMR

7

Mixer

Routing Matrix



My Country 'Tis of Thee [modified] [proj 1/2] - REAPER v5.18 - EVALUATION LICENSE

File Edit View Insert Item Track Options Actions Help [Change track selection]

[44.1kHz 24bit WAV : 2/2ch 1024spl/s ~43/94ms WaveOut]

\*My Country 'Tis of Thee.RPP \*Solon Pierce Piano Session.RPP

CLOSE MIC

CML

CMR

ROOM MIC

RML

RMR

1.1.00 257.1.00 513.1.00 769.1.00 1025.1.00 1281.1.00 1537.1.00 1793.1.00 2049.1.00 2305.1.00 2561.1.00 2817.1.00 3073.1.00 3329.1.00 3585.1.00

0-00.000 8-32.000 17-04.000 25-36.000 34-08.000 42-40.000 51-12.000 59-44.000 1-08-16.000 1-16-48.000 1-25-20.000 1-33-52.000 1-42-24.000 1-50-56.000 1-59-28.000

FX: Track 1 "CLOSE MIC" (folder)

FX Edit Options

VST: ReaEQ (Cockos)

stock - Close Mic Acoustic

Param 2 in 2 out UI

ReaEQ

Gain:

0.0

1 2 3 4 5 6

1

2

3

4

5

Enabled

Type: High Pass

Log-scale automated frequencies

Frequency (Hz): 129.1 C3

Gain (dB): 0.0

Bandwidth (oct): 2.00

Add band Remove band Reset defaults Show tabs Show grid Show phase

0.2%/0.2% CPU 0/0 spl/s

745.3.95 / 24:49.473 [Stopped]

Rate: 1.0 BPM 120 4/4 Selection: 1.4.55 208.3.27 204.2.72

MASTER

CLOSE MIC

CML

CMR

ROOM MIC

RML

RMR

Mixer Routing Matrix

10:20 PM 5/3/2017





Playback Rate: 1.00000





Project browser and track list showing tracks 01, 02, and 03. Track 01 is selected.

Timeline showing time markers from 1.1.00 to 273.1.00.

Waveform view showing audio data for Track 01, Track 02, and Track 03.

Mixer section showing volume and pan controls for tracks 01, 02, and 03. The Master track is also visible.

FX: Track 1 "I" window showing the ReaEQ plugin interface.

FX Edit Options:

- ☒ VST: ReaEQ (Cockos)
- ☒ VST: ReaEQ (Cockos)
- ☒ VST: ModernLimiter (Antress)

ReaEQ settings:

- No preset
- Gain: 0.0
- Frequency (Hz): 100.0
- Gain (dB): 0.0
- Bandwidth (oct): 2.00
- Type: Low Shelf
- Log-scale automated frequencies: ☒

Buttons: Add band, Remove band, Reset defaults, Show tabs, Show grid, Show phase.

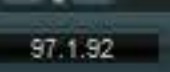
Waveform view showing audio data for Track 01, Track 02, and Track 03. The waveform for Track 01 is highlighted in green.





3 03 Track      

4.2.25 / 0:06.624

☒ Mixer ☒ Routing Matrix